

## CLAIMS

*Sub A17*

1. A method of supporting communications among a plurality of  
2 communications terminals, comprising the steps of:  
3 receiving an audio signal from a first communications terminal;  
4 processing the audio signal according to desired acoustical procedures;  
5 formulating one or more mixes of the audio signal, wherein each mix is  
6 associated with a communications terminal; and  
7 sending each of the formulated mixes to its associated communications  
8 terminal.

1 2. The method of claim 1, further comprising the steps of:  
2 decoding the received audio signal; and  
3 encoding at least one of the formulated mixes prior to sending the mix to its  
4 associated communications terminal.

1 3. The method of claim 1, wherein the processing step comprises the step of:  
2 processing the audio signal responsive to room model information held in a  
3 room model associated with the first communications terminal.

1 4. The method of claim 3, wherein the room model information includes data  
2 representative of acoustic properties of the first communications terminal.

1 5. The method of claim 3, wherein the room model information includes data  
2 representative of acoustic properties of an environment surrounding the first  
3 communications terminal.

1 6. The method of claim 3, further comprising the steps of:  
2 sampling data representative of a formulated mix sent to a second  
3 communications terminal; and

storing the sampled data in a room model associated with the second communications terminal.

7. The method of claim 1, further comprising the steps of:  
decoding the received audio signal;  
processing the decoded audio signal responsive to room model information  
associated with the first communications terminal; and  
encoding at least one of the formulated mixes prior to sending the mix to its  
associated communications terminal.

8. The method of claim 7, further comprising the steps of:  
sampling data representative of a formulated mix sent to a second  
communications terminal; and  
storing the sampled data in a room model associated with the second  
communications terminal.

9. The method of claim 8, further comprising the step of:  
decoding the encoded formulated mix associated with the second  
communications terminal;  
wherein the sampling step samples the decoded encoded formulated mix  
associated with the second communications terminal.

10. The method of claim 1, wherein the processing step comprises the step of:  
performing acoustic echo cancellation on the audio signal.

11. The method of claim 1, wherein the processing step comprises the step of: performing automatic gain control on the audio signal.

12. The method of claim 1, wherein the processing step comprises the step of:  
performing noise reduction on the audio signal.

13. The method of claim 1, wherein the processing step comprises the step of:  
2 determining an amount of processing power to allocate to processing the audio  
3 signal responsive to a number of communications terminals engaged in  
4 the communications session.

14. The method of claim 1, wherein the processing step comprises the step of:  
2 determining an amount of processing power to allocate to processing the audio  
3 signal responsive to characteristics of the audio signal.

15. The method of claim 1, wherein the sending step comprises the step of:  
2 sending time tags including sequencing information with each of the  
3 formulated mixes.

16. The method of claim 15, further comprising the step of:  
2 receiving time tags with the audio signal;  
3 wherein the processing step aligns the audio signal responsive to sent and  
4 received time tags.

17. A communications system for supporting a communications session  
2 among a plurality of terminals, wherein each terminal has a microphone and a speaker,  
3 the communications system comprising:  
4 a communications support module (CSM) for receiving audio signals from the  
5 microphones of one or more of the plurality of terminals engaged in  
6 the communications session;  
7 a signal processing (SP) module for performing signal processing on the  
8 received audio signals under direction of the CSM to produce  
9 processed audio signals;  
10 a mixing module for mixing the processed audio signals under direction of the  
11 CSM to produce one or more output mixes, wherein each output mix is  
12 associated with a terminal engaged in the communications session; and

13                   a module for sending each output mix to its associated terminal for output on  
14                   the terminal's speaker.

1           18. The system of claim 17, further comprising:  
2            a room model module for holding one or more room models associated with  
3            one or more of the terminals, wherein a room model holds room model  
4            information, and wherein the SP module performs signal processing on  
5            the received audio signals responsive to the room model associated  
6            with the terminal from which the audio signal was received.

1           19. The system of claim 17, wherein the SP module performs acoustic echo  
2           cancellation on the received audio signals.

1           20. The system of claim 17, wherein the SP module performs automatic gain  
2           control on the received audio signals.

1           21. The system of claim 17, wherein the SP module performs noise reduction  
2           on the received audio signals.

1           22. The system of claim 17, wherein the received audio signals are encoded  
2           and further comprising:  
3            a decoding module for decoding the received audio signals; and  
4            an encoding module for encoding at least one output mix prior to the output  
5            mix being sent to its associated terminal.

1           23. The system of claim 22, wherein the decoding module is adapted for  
2           decoding at least one encoded output mix to produce a decoded encoded output mix,  
3           wherein the decoded encoded output mix is used by the SP module to perform acoustic  
4           echo cancellation.

1        24. The system of claim 17, wherein the CSM determines an amount of SP  
2 module processing resources to allocate to processing particular received audio signals  
3 responsive to a load on the communications system.

1        25. The system of claim 17, wherein the CSM determines an amount of SP  
2 processing resources to allocate to processing particular received audio signals responsive  
3 to characteristics of the received audio signals.

1        26. A system for supporting a communications session among a plurality of  
2 communications terminals, the system comprising:

3        a receiving module for receiving audio signals from the plurality of  
4        communications terminals;  
5        a room model module for storing room models associated with one or more of  
6        the plurality of communications terminals, the room models holding  
7        room model information;

8        a signal processing (SP) module for performing SP on the audio signals  
9        received from the plurality of communications terminals responsive to  
10      the room models associated with the communications terminals from  
11      which the audio signals were received, to produce processed audio  
12      signals; and

13      a sending module for sending the processed audio signals to the plurality of  
14      communications terminals.

1        27. The system of claim 26, further comprising:

2        a mixing module for mixing selected ones of the processed audio signals to  
3        produce one or more mixes, wherein the sending module sends the one  
4        or more mixes to selected ones of the communications terminals.

1        28. The system of claim 26, wherein the SP module is adapted to perform  
2        acoustic echo cancellation on the received audio signals.

1           29. The system of claim 26, wherein the SP module is adapted to perform  
2 automatic gain control on the received audio signals.

1           30. The system of claim 26, wherein the SP module is adapted to perform  
2 noise reduction on the received audio signals.

1           31. The system of claim 26, wherein the received audio signals are encoded  
2 and further comprising:

3           an encoding/decoding module for decoding the audio signals received by the  
4           receiving module and for encoding the processed audio signals sent by  
5           the sending module.

1           32. The system of claim 26, wherein the SP resources are allocated to the  
2 audio signals responsive to a total amount of available SP resources.

1           33. The system of claim 26, wherein the SP resources are allocated to the  
2 audio signals responsive to characteristics of the received audio signals.

1           34. A method of sharing signal processing resources to support  
2 communications among a plurality of communications terminals, comprising the steps of:  
3           maintaining a central pool of signal processing resources;  
4           determining whether a communications terminal of the plurality of  
5           communications terminals is active;  
6           responsive to determining that the communications terminal is active,  
7           allocating a portion of the pool of signal processing resources to  
8           processing a signal from the communications terminal; and  
9           responsive to determining that an audio signal is not active, deallocating a  
10           portion of the pool of signal processing resources from processing the  
11           signal from the communications terminal.

1        35. The method of claim 34, wherein the step of determining whether a  
2        communications terminal of the plurality of communications terminals is active  
3        comprises the steps of:  
4                receiving data packets from the communications terminal, the data packets  
5                forming the signal from the communications terminal; and  
6                analyzing the data packets to determine whether an audio signal is originating  
7                at the communications terminal.

1        36. The method of claim 34, wherein the allocating step comprises the steps  
2        of:  
3                determining an amount of signal processing resources available in the pool;  
4                and  
5                determining a number of communications terminals of the plurality of  
6                communications terminals that are producing audio signals;  
7                wherein the portion of the pool allocated to processing the signal from the  
8                communications terminal is determined responsive to the amount of  
9                signal processing resources available in the pool and the number of  
10          communications terminals that are producing audio signals.

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